

**Changes to the Specification**

Please insert the following as the first paragraph of page 1:

This application is a continuation of application no. 09/792,105, filed February 22, 2001, now Patent No. 6,628,774, which is a continuation of application no. 09/197,386, filed November 20, 1998, now Patent No. 6,400,818, which is a continuation-in-part of application no. 08/768,382, filed December 17, 1996, now Patent No. 5,917,899 and a continuation-in-part of application no. 08/766,598, filed December 12, 1996, now Patent No. 5,987,111.

Please replace the paragraph on page 11, lines 13-23 with the following amended paragraph:

When the original called party ID is present, it is determined if the call is an access to virtual networks call at step 158. When the call is not an access to virtual networks call, then it is ~~determine~~ determined if the call is a toll aggregation call at step 160. When the call is not the toll aggregation call, an error has occurred and a terminating announcement is played at step 156. When the call is the toll aggregation call, an analyze response message is sent to the hub SSP at step 162. The hub SSP routes the call to a preferred inter-exchange carrier at step 164, which ends call processing at step 166. The toll aggregation processing 158-166 is explained in detail with respect to FIGs. 6-7.

Please replace the paragraph on page 12, lines 4-21 with the following amended paragraph:

FIG. 5 is a flow chart of an embodiment of the steps performed by a second hub service switching point, the switching control point and a second service switching point

according to the invention. The second hub SSP receives the customized dialing plan (CDP) abbreviated dial code and trigger on the code at step 180. The second hub SSP then sends an information analyzed query to the SCP at step 182. The query includes the abbreviated dial code. The SCP determines if the call is in-network at step 184. When the call is not in-network an error has occurred in the service at step 186. When the call is in-network, the SCP then sends an analyzed route message containing a routing instruction at step 188. The hub SSP then routes the call to a second hub SSP at step 190 based on the routing instruction. The second hub SSP also sends an IAM to the second SSP (second service switching point) at step 192 containing a translated routing number. The translated routing number is the directory number associated with the abbreviated dialing code. The second SSP then routes the call to the called ~~part~~ party at step 194, which ends call processing at step 196.

Please replace the paragraph on page 13, lines 8-13 with the following amended paragraph:

When the access code is present, an information analyzed query is sent to the SCP at step 212. The SCP then determines if the call is restricted at step 214. When the call is restricted, a restricted call response message is sent to the SSP at step 216. The SSP then plays the terminating announcement that the call is not ~~authorize~~ authorized at step 218, which ends the processing at step 220.